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WHAT IS CLAIMED IS:

1. A communications apparatus comprising:

an encoder for encoding a signal;

a code compression unit, coupled to the encoder, for compressing the encoded signal using a lossless scheme and a lossy scheme; and

- a memory, coupled to an output of the code compression unit, for storing the compressed encoded signal.
 - 2. The apparatus of claim 1 further comprising:

a code decompression unit, coupled to the memory, for decompressing the stored signal using a lossless scheme and a lossy scheme; and

- a decoder, coupled to the code decompression unit, for decoding the decompressed signal.
 - 3. The apparatus of claim 2 wherein the quality of the signal decompressed using the lossy scheme is improved by changing weighting factors and a tilt factor in a post filter.
- 4. The apparatus of claim 1 wherein the lossless scheme is used to compress parameters of the encoded signal having high inter-frame redundancy.

- 5. The apparatus of claim 4 wherein the parameters of the encoded signal having high inter-frame redundancy includes coefficients of a long term filter and codebook gains.
- 6. The apparatus of claim 1 wherein the lossy scheme is used to compress some parameters of the encoded signal having low inter-frame redundancy.
- 7. The apparatus of claim 6 wherein the parameters of the encoded signal having low inter-frame redundancy that are compressed include fixed codebook indices.
- 8. The apparatus of claim 6 wherein the parameters of the encoded signal having low inter-frame redundancy that are not compressed include adaptive codebook indices.
- 9. The apparatus of claim 1 further comprising a switch that enables an encoded signal received by a receiver to be compressed by the code compression unit and stored in the memory.
- 10. The apparatus of claim 2 further comprising a switch that enables the stored signal to be decompressed by the decompression unit and output from a transceiver.

- 11. The apparatus of claim 1 further comprising an operator interface unit.
- 12. The apparatus of claim 1 wherein the apparatus is a mobile telephone or a communication device.

13. A method for compressing a signal comprising the steps of: converting the signal to a digital signal; encoding the digital signal;

compressing, within a compression unit, the encoded signal using a lossless scheme and a lossy scheme; and

storing the compressed encoded signal in a memory coupled to an output of the compression unit.

14. The method of claim 13 further comprising the steps of:

decompressing, within a decompressing unit, the stored signal using a lossless scheme and a lossy scheme;

decoding, within a decoder, the decompressed signal; and outputting the decoded signal.

- 15. The method of claim 14 wherein the quality of the signal decompressed using the lossy scheme is improved by changing weighting factors and a tilt factor in a post filter of the decoder.
- 16. The method of claim 13 wherein the lossless scheme is used to compress parameters of the encoded signal having high inter-frame redundancy.

- 17. The method of claim 16 wherein the parameters of the encoded signal having high inter-frame redundancy include coefficients of a long term filter and codebook gains.
- 18. The method of claim 13 wherein the lossy scheme is used to compress some parameters of the encoded signal having low inter-frame redundancy.
- 19. The method of claim 18 wherein the parameters of the encoded signal having low inter-frame redundancy that are compressed include fixed codebook indices.
- 20. The method of claim 18 wherein the parameters of the encoded signal having low inter-frame redundancy that are not compressed include adaptive codebook indices.

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21. A method of improving quality of a lossy-compressed signal comprising the steps of:

performing a lossy compression of an uncompressed signal to yield a lossy-compressed signal;

performing a transform of the uncompressed signal from time domain to frequency domain;

decompressing the lossy-compressed signal;

performing a transform of the decompressed lossy-compressed signal from time domain to frequency domain;

comparing an absolute value of the transformed uncompressed signal to the absolute value of the transformed decompressed lossy-compressed signal;

adjusting weighting factors and a tilt factor until a minimal difference between the absolute values of the transformed signals is reached; and

applying the adjusted weighting factors and the adjusted tilt factor to
the decompressed lossy-compressed signal.

- 22. The method of claim 21 wherein the transforms are performed using short time Fourier transforms.
- 23. The method of claim 21 wherein the method is performed in an AMR codec.

- 24. The method of claim 21 wherein the method is performed in an EFR codec.
- 25. The method of claim 21 further comprising the step of performing a subjective listening test to confirm the adjusted factors.

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26. An apparatus for improving quality of a lossy-compressed signal comprising:

a code compression unit adapted to lossy-compress an uncompressed signal;

a code decompression unit adapted to decompress the lossy-compressed signal; and

a processor adapted to:

perform a transform of the uncompressed signal and of the decompressed lossy-compressed signal from time domain to frequency domain;

compare an absolute value of the transformed uncompressed signal to an absolute value of the transformed decompressed lossy-compressed signal; and adjust weighting factors and a tilt factor until a minimal difference between the absolute values of the transformed signals has been reached.

- 27. The apparatus of claim 26 further comprising a post filter adapted to apply the adjusted weighting factors and the adjusted tilt factor to the decompressed lossy-compressed signal.
- 28. The apparatus of claim 27 wherein the apparatus comprises part of an EFR codec.
- 29. The apparatus of claim 27 wherein the apparatus comprises part of an AMR codec.

30. A method of sorting parameters of an encoded speech signal for compression comprising the steps of:

determining a degree of inter-frame redundancy of each of the parameters;

lossy compressing a first portion of the parameters, the first portion having relatively low inter-frame redundancy; and

losslessly compressing a second portion of the parameters, the second portion having relatively high inter-frame redundancy.

- 31. The method of claim 30 further comprising the step of not compressing a third portion of the parameters, the third portion of the parameters being selected according to pre-determined criteria irrespective of inter-frame redundancy.
- 32. The method of claim 30 wherein the degree of inter-frame redundancy of each of the parameters is determined by statistical analysis.
- 33. The method of claim 30 the second portion includes coefficients of a long term filter and codebook gains.
- 34. The method of claim 30 wherein the first portion includes fixed codebook indices and adaptive codebook indices.

35. A method for decompressing a signal comprising the steps of:

decompressing, within a decompressing unit, a compressed encoded digital signal using a lossless scheme and a lossy scheme;

decoding, within a decoder, the decompressed signal; and outputting the decoded signal.

- 36. The method of claim 35 wherein the quality of the decompressed signal is improved by changing weighting factors and a tilt factor in a post filter of the decoder.
- 37. The method of claim 35 further comprising the step of losslessly interframe redundancy.
- 38. The method of claim 37 wherein the parameters of the encoded digital signal having high inter-frame redundancy include coefficients of a long term filter and codebook gains.
- 39. The method of claim 35 further comprising the step of lossy compressing some parameters of an encoded digital signal, the parameters having low inter-frame redundancy.

- 40. The method of claim 39 wherein the parameters of the encoded signal having low inter-frame redundancy include fixed codebook indices.
- 41. The method of claim 39 wherein the parameters of the encoded signal having low inter-frame redundancy include adaptive codebook indices.